

B. REJECTION OF INDEPENDENT CLAIMS 11 AND 19 AND CLAIMS 12, 13, 21, 22, 36 AND 38 DEPENDENT THEREFROM UNDER 35 USC 103 BASED ON CHEN IN VIEW OF FRANGER

The Final Rejection states that Chen discloses (1) a coder that performs error correction coding (ECC) of input data including a plurality of bits; and (2) an interleaver that interleaves the bits coded by the coder, but Chen lacks a rate matcher that comprises a repeater and a puncturer, wherein the rate matcher alternatively selects between (i) employing the repeater to repeat a part of the bits interleaved by the interleaver and (ii) employing the puncturer to puncture a part of the bits interleaved by the interleaver.

The Final Rejection relies on Franger for a disclosure of a rate matcher that includes a repeater and a puncturer.

The Final Rejection asserts "Therefore, it would have been obvious to those skilled in the art, at the time the invention was made, to incorporate the rate matching including a repeater and a puncturer as taught by Franger et al into the invention of Chen et al to repeat or puncture a part of the bits interleaved by the interleaver to provide flexible and efficient method for source data rate matching."

The Applicants respectfully submit that Chen and Franger are deficient to render obvious the present claims for the following reasons.

Chen's FIG. 1 shows a system wherein the symbols from a convolutional encoder 114 are interleaved in an interleaver 115 and then applied to modulator 116. As noted by the final Rejection, Chen does not disclose rate matching.

Franger discloses both "rate-matching" and "interleaving" although he *does not disclose the order of these operations*. Specifically, Franger does not teach or suggest that puncturing (rate matching) is performed after interleaving. Further, the Applicant submits that, in view of the Hagenauer article cited by Frenger ("Rate-Compatible Punctured Convolutional Codes (RCPC Codes) and their Applications"), a copy of which is attached, Frenger's "interleaving" would be performed after the "rate-matching." This paper by Hagenauer is a fundamental concept of the RCPC encoding of Frenger. The RCPC code used in Frenger was originally provided by Hagenauer. Significantly, in Fig. 4 of Hagenauer, it is clear that the "interleaving" is performed after RCPC encoding which includes "puncturing." There is nothing in Franger that would have suggested altering this order taught by Hagenauer.

Accordingly, it is submitted that a combination of Chen and Franger would not have suggested the present claimed invention because:

- (1) Franger does not disclose or suggest "puncturing" performed after "interleaving."
- (2) The Hagenauer article which was cited in Frenger discloses that "interleaving" is performed after RCPC encoding (which includes puncturing).
- (3) Franger lacks anything that would suggest altering the order taught by Hagenauer.
- (4) The "puncturing" of Frenger is performed within a "coding" scheme (RCPC encoding).
- (5) Chen discloses "interleaving" performed after "coding";
- (6) Thus, according to (1)-(5), even if Chen and Frenger were combined, the "puncturing" would be performed within the coding before "interleaving."

Finally, the "repetition" of Frenger is merely a reference point to prove the higher performance of "puncturing"; thus, Frenger does not disclose or suggest the subject matter recited in the present claims directed to performance of rate matching including "alternatively selecting between (i) repeating a part of the interleaved bits and (ii) puncturing a part of the interleaved bits."

Accordingly, it is submitted that claims 11, 19 and 23, and all claims dependent therefrom, recite allowable subject matter.

C. REJECTION OF INDEPENDENT CLAIMS 32, 35 AND 40 AND CLAIMS 20, 37, 39, 42, AND 44 DEPENDENT THEREFROM UNDER 35 USC 103(A) AS UNPATENTABLE OVER CHEN ET AL. IN VIEW OF FRANGER ET AL. IN VIEW OF THE APPLICANT'S ADMITTED PRIOR ART

Chen and Franger have been discussed and distinguished above. The Applicants' admitted prior art merely discloses a coding device in which, first, the number of bits is increased, and then, the bits are rearranged in an interleaver.

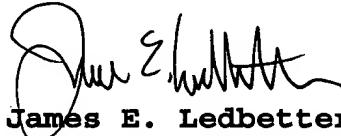
However, claim 35 is directed to receiving data transmitted by the method of claim 22, including, first, alternatively selecting between repeating and puncturing bits, and, then, deinterleaving data. Claims 32 and 40 are directed to, first, receiving data transmitted by the methods of claim 31 and 34, respectively, then, puncturing bits, and then, deinterleaving the data. Thus, these claims are directed to, on the reception side, repeating or puncturing and then de-interleaving. This is opposite to the order of the admitted prior art and is not taught or suggested by the combined teachings of Chen, Franger and the admitted prior art.

Accordingly, it is submitted that independent claims 32, 35 and 40, and all claims dependent therefrom, recite allowable subject matter.

A notice of allowance is respectfully solicited.

If any issues remain that may best be addressed through a telephone communication, the examiner is requested to telephone the undersigned at the local Washington, D.C. telephone number listed below.

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Rate-Compatible Punctured Convolutional Codes (RCPC Codes) and their Applications

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Abstract—The concept of punctured convolutional codes is extended by puncturing a low rate $1/N$ code periodically with period P to obtain a family of codes with rate $P/(P+1)$ where 1 can be varied between 1 and $(N-1)/P$. A rate-compatibility restriction on the puncturing tables ensures that all code bits of high rate codes are used by the lower rate codes. This allows transmission of incremental redundancy in ARQ/FEC schemes and continuous rate variation to change from low to high error protection within a data frame. Families of RCPC codes with rates between $8/9$ and $1/4$ are given for memories M from 3 to 6 (8 to 64 trellis states) together with the relevant distance spectra. These codes are almost as good as the best known general convolutional codes of the respective rates. It is shown that the same Viterbi decoder can be used for all RCPC codes of the same M by controlling the metric memory access through the puncturing rule, the soft decision and the channel state information (CSI). The performance of RCPC codes on Gaussian and fading channels is shown under various quantization and CSI conditions. The application of RCPC codes to hybrid ARQ/FEC schemes is discussed for Gaussian and Rayleigh fading channels using CSI to optimize throughput. If a source coder supplies a source significance information (SSI) indicating the relative importance of bits, a flexible unequal error protection scheme can be designed with RCPC codes which minimizes the redundancy overhead.

I. INTRODUCTION

THE design of an error correction coding system usually consists of selecting a fixed code with a certain rate and correction capability matched to the protection requirement of all the data to be transmitted and adapted to the average or worst channel conditions to be expected [1]. In many cases, however, one would like to be more flexible because the data to be transmitted have different error protection needs and the channel is time varying or has insufficiently known parameters. Consequently, flexible channel encoding and an adaptive decoder are required. As Fig. 1 shows, the information to be transmitted might carry source significance information (SSI) indicating different protection requirements. Examples include control and signaling bits in a data stream, most significant bits in a PCM word, or side information like quantizer slopes, or filter coefficients in more sophisticated voice encoding schemes [2]. On the other hand, the channel characteristics or the channel state might vary considerably, as encountered in mobile or multipath radio transmission, in a jamming environment, during rain fading, or in HF transmission. This is indicated in Fig. 1 by the channel state information (CSI). In rare cases the instantaneous CSI is available at the encoder where code adaptation could take place. Mostly the receiver only can use a CSI-like fading

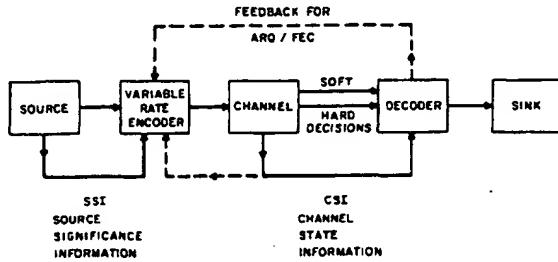


Fig. 1. Coded transmission scheme with source significance information (SSI) and channel state information (CSI).

depth, noise level variation, short-term signal loss, or jammer activity. The CSI can significantly improve decoder performance together with soft decisions at the receiver. Whenever a return channel is available, the CSI can be indirectly relayed to the transmitter by asking for a retransmission as employed in automatic repeat request (ARQ) systems. Such ARQ systems can also be combined with forward error correction (FEC) to yield type I or type II ARQ/FEC hybrid schemes [3], [4]. The latter schemes nicely match the average channel rate or the throughput with the channel conditions and some of them use different FEC codes for repeated transmission attempts. The scenarios shown in Fig. 1 require variable codes adapted to the source and channel needs. We wish to change the code rate, i.e., the number of check bits, and hence the correction power of the code during transmission of an information frame according to source and channel needs. For practical purposes, we would like to have not just switching between a set of encoders and decoders, but one encoder and one decoder which can be modified without changing their basic structure. This can be achieved by not transmitting certain code bits, namely, by puncturing the code. Mandelbaum [5] was the first to propose punctured codes for transmitting redundancy in incremental steps by using Reed-Solomon codes. In order to accommodate soft decisions and CSI at the receiver, a maximum likelihood decoder is required. This motivates the use of convolutional codes and the Viterbi algorithm for decoding.

Punctured convolutional codes were first introduced by Cain, Clark, and Geist [6] mainly for the purpose of obtaining simpler Viterbi decoding for rate K/N codes with two branches arriving at each node instead of 2^K branches. They obtained codes of rate $2/3$ and $3/4$ by puncturing rate $1/2$ codes. These punctured codes were almost as good as the best known codes. Some of the good codes used the same basic rate $1/2$ generators. Later, Yasuda *et al.* [7], [8] found a family of $(N-1)/N$ codes by puncturing $1/2$ codes for N up to 14 , and built selectable rate encoders and Viterbi decoders using soft decisions.

In this paper, the concept of punctured convolutional codes is modified for the generation of a family of codes by adding a rate-compatibility restriction to the puncturing rule. The restriction implies that all the code bits of a high rate punctured

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code are used by the lower rate codes; or in other words, the high rate codes are embedded into the lower rate codes of the family. If the higher rate codes are not sufficiently powerful to decode channel errors, only supplemental bits which were previously punctured have to be transmitted in order to upgrade the code. Furthermore, since codes are compatible, rate variation within a data frame is possible to achieve unequal error protection. In Section II, the necessary definitions and the performance criterion for the class of rate compatible punctured convolutional codes (RCPC codes) are given. Section III reports the results of a computer search for families of RCPC codes of rate $P/(P+1)$, with rates between 1/4 and 8/9. The analytically calculated and simulated performance of some codes on Gaussian, Rayleigh, and Rice fading channels is shown in Section IV. Modified type II ARQ/FEC schemes with RCPC schemes shown in Section V allow a minimum number of transmitted bits as well as soft decision and CSI adaptive decoding and compare favorable to previous known schemes. Unequal error protection with RCPC codes is discussed in Section VI.

II. DEFINITION AND PERFORMANCE CRITERION OF RCPC CODES

In order to explain rate-compatible punctured codes, we start with the example of Fig. 2 where a rate $R = 1/2$ convolutional code with memory $M = 2$ is punctured periodically with period $P = 4$. We describe the binary information symbols and the binary code symbols x_{ij} and x_{ij} by values ± 1 . A zero in the puncturing table means that the code symbol is not to be transmitted. In the upper example of Fig. 2, the fourth bit of the upper branch and the second and third bit of the lower branch are not transmitted. The puncturing table can be viewed as a modulo P rule for multiplexing the two streams of code bits. Instead of transmitting $2 \cdot P = 2 \cdot 4$ only $P + 1 = P + 1 = 5$ bits are transmitted per $P = 4$ information bits. Therefore, we have generated a code with rate $R = 4/5$ by using only $P + 1$ '1' 's instead of $2P$ '1' 's in the puncturing table, which can be described by the $N \times P$ matrix

$$a(1) = \begin{pmatrix} 1 & 1 & 1 & 0 \\ 1 & 0 & 0 & 1 \end{pmatrix}.$$

Suppose the code rate with 4/5 and with the puncturing table $a(1)$ is not powerful enough to correct the channel errors. A more redundant and therefore more powerful code with lower rates 4/6, 4/7, or 4/8 would be necessary. Instead of transmitting all the code bits of a completely different low rate code, the lower rate code should utilize the bits already transmitted. Then only additional incremental redundancy bits have to be transmitted. Additional '1' 's in the puncturing tables of the lower rate codes can therefore be placed only where zeros appeared in the puncturing matrix of the previous higher rate code, for example,

$$a(2) = \begin{pmatrix} 1 & 1 & 1 & 0 \\ 1 & 1 & 0 & 1 \end{pmatrix}, \quad a(3) = \begin{pmatrix} 1 & 1 & 1 & 1 \\ 1 & 1 & 0 & 1 \end{pmatrix},$$

$$a(4) = \begin{pmatrix} 1 & 1 & 1 & 1 \\ 1 & 1 & 1 & 1 \end{pmatrix}.$$

Thus, for $l = 1$ to 4, we get a compatible family of codes derived from the mother code 1/2 with rates 4/5, 4/6, 4/7, 4/8 which use only incremental redundancy. The question, of course, is whether all these codes in the family are good and whether noncatastrophic codes exist. This will be discussed in Section III. On the receiving side the decoder using the Viterbi algorithm (VA) has to know the current puncturing rule a . The VA receives real values y_{ij} when full soft decision is used. Alternatively, the decoder can use quantized values of y_{ij} . The VA finds the path with the maximum likelihood metric for

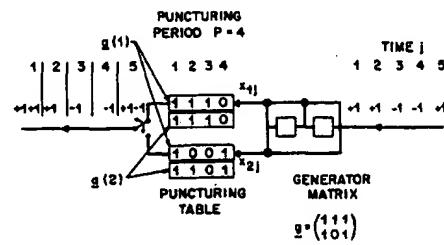


Fig. 2. Example of a punctured convolutional code with two rate compatible puncturing tables.

statistically independent y_{ij} by calculating

$$\max_m \sum_{j=1}^J \lambda_j \quad (1)$$

where the metric increment is

$$\lambda_j = \sum_{i=1}^N a_{ij} x_{ij}^{(m)} y_{ij} \quad (2)$$

and $a_{ij, P} = a_{ij}$ due to periodic puncturing. $a_{ij} = 0$ means that this $x_{ij}^{(m)}$ has not been transmitted and y_{ij} is not available. Actually, it is not necessary to insert any dummy data at the receiver as suggested in [7]. The Viterbi algorithm operates on the trellis of the $1/N$ mother code with two branches leaving at each state. Equation (2) simply means that for $a_{ij} = 1$ the metric ROM is accessed using the address $x_{ij}^{(m)} y_{ij}$ to add a metric increment and for $a_{ij} = 0$ it is not accessed.

For a fading channel, the transmitted signal is multiplied by a fading factor a_f . If interleaving is assumed, a_f is an independent positive real random variable with density $p(a_f)$, i.e., Rayleigh or Rice. If this factor can be estimated at the receiver and is used as a CSI the ML-metric for coherent transmission is [9]

$$\lambda_j = \sum_{i=1}^N a_{ij} a_{ij, P} x_{ij}^{(m)} y_{ij}. \quad (3)$$

As expected, a deep fade ($a_{if} \rightarrow 0$) has the same effect as puncturing ($a_{ij} = 0$).

A. General Definition of RCPC Codes

A family of RCPC codes is described by the mother code of rate $R = 1/N$ and memory M having the generator tap matrix

$$g = \overset{\leftarrow M+1 \rightarrow}{\underset{\leftarrow \downarrow \rightarrow}{\overset{\uparrow}{N}}} \left(g_{ik} \right) \quad (4)$$

with the tap connections $g_{ik} \in \{0, 1\}$ where a 1 represents a connection from the k th shift register stage to the i th output. Together with N , the puncturing period P determines the range of code rates

$$R = \frac{P}{P+1} \quad l = 1, \dots, (N-1)P \quad (5)$$

between $P/(P+1)$ and $1/N$. The RCPC codes are punctured codes of the mother code with puncturing matrices

$$a(l) = \overset{\leftarrow P \rightarrow}{\underset{\leftarrow \downarrow \rightarrow}{\overset{\uparrow}{N}}} \left(a_{ij}(l) \right) \quad (6)$$

with $a_{ij}(l) \in \{0, 1\}$ where 0 implies puncturing.

The rate-compatibility restriction implies the following rule:

$$\text{if } a_{ij}(l_0) = 1 \text{ then } a_{ij}(l) = 1 \text{ for all } l \geq l_0 \geq 1 \quad (7a)$$

or equivalently

$$\text{if } a_{ij}(l_0) = 0 \text{ then } a_{ij}(l) = 0 \text{ for all } l \leq l_0 \leq (N-1)P-1. \quad (7b)$$

B. Criterion of Goodness for RCPC Codes

With Viterbi decoding the usual optimality criterion is a large free distance d_{free} , a small number of paths a_d , and a small information error weight c_d on all paths with $d \geq d_{\text{free}}$. More specifically, one has Viterbi's upper bound an error event probability [13]

$$P_E \leq \frac{1}{P} \sum_{d=d_{\text{free}}}^{\infty} a_d P_d \quad (8)$$

and for the bit error probability

$$P_b \leq \frac{1}{P} \sum_{d=d_{\text{free}}}^{\infty} c_d P_d. \quad (9)$$

P_d is the probability that the wrong path at distance d is selected. The so-called distance spectra $\{a_d\}$ and $\{c_d\}$, which should be as small as possible, depend only on the code, namely, on N , M , P , g , and $a(l)$. Due to the time-varying nature of the RCPC codes, P different starting points for diverging paths have to be considered to obtain the total numbers a_d and c_d . Since RCPC codes are a subclass of punctured codes, which in turn is a subclass of the general K/N codes, it has still to be shown that the RCPC restriction is not too severe to prevent "good" codes from being located in this subclass. A degree of freedom can be gained by increasing the period P . RCPC codes constitute a specific class of time-varying codes with a fixed generator but periodically time-varying puncturing. We know that some time-varying codes are better than fixed codes [10] and some random coding theorems hold only for time-varying codes. Therefore, some improvement may be possible by using a higher period P . Note that P has to be sufficiently large if codes of specific rates are to occur in the same RCPC family. The complexity of the VA decoder of an RCPC code in terms of add, compare and select operations per decoded bit is $2 \cdot 2^M$, whereas a general K/N code requires $(2^k 2^M)/K$ such operations where M is the total encoder memory. At the RCPC decoder, an $N \cdot P$ ambiguity has to be resolved in the incoming data stream and the decision depths of punctured codes are generally longer [8]. The last disadvantage is not severe, when as in most applications, data frames are transmitted with sync words and proper termination of short frames.

III. RESULTS OF THE SEARCH FOR GOOD RCPC CODES

No constructive method is known for determining the generator matrix g and the puncturing matrices $a(l)$ for a RCPC code family. Therefore a computer search has been performed under some restrictions. For distances $d \geq d_{\text{free}}$, a_d , c_d , and the path length distribution has been determined by a stack algorithm similar to Cedervall's algorithm [11].

The basic steps to find a family of good RCPC codes are:

Step 1: Select mother code with rate $1/N$, memory M , and generator g . Select puncturing period P . For $l = l_{\text{max}} = (N-1)P$ set $a(l) = (1)$.

For $l = (N-1)P$ to 2 perform steps 2-4.

Step 2: Amongst the elements $a_{ij}(l)$ which are "1" select one and set it to "0" [compatibility rule (7b)]. Observe that cyclic column permutations give the same code performance.

Step 3: Determine d_{free} and c_d for $d \geq d_{\text{free}}$.

Step 4: Repeat steps 2 and 3 to find $a(l-1)$ with maximal d_{free} and minimal c_d for $d = d_{\text{free}}, d_{\text{free}} + 1$. If only catastrophic codes can be found, use next best code at level l .

The results of the computer search for the $M = 4$ code family are given in some detail in Table I, whereas for the other codes of memory $M = 3, 5, 6$, only $a(l)$ and the c_d values are given in Table II. The codes have a period $P = 8$ with a subperiod of $P = 4$ and are punctured mostly in steps $\Delta l = 2$ in such a way as to arrive at the best known code of rate $1/2$. This implies that the mother codes of rate $1/3$ or $1/4$ are not necessarily the best known code of this rate. This is the case for $M = 4$ and 6 where one path exists at a distance which is one less than the best known free distance [3].

Table III shows a comparison of the most important distance parameters of RCPC codes, punctured codes, and general K/N codes for rates where results are available. This and other comparisons show that RCPC codes are almost as good as the other codes. By increasing the period P one can even obtain slightly better codes as shown for $M = 3$ and rate $2/3$ code. $P = 4$ gives a greater variety in selecting the puncturing positions. In this case, c_d/P for $d = d_{\text{free}}, \dots, d_{\text{free}} + 5$ is less than the corresponding numbers for the code with $P = 2$ and therefore a better BER performance can be expected over the whole SNR range. This result is a further confirmation that time-varying codes are better than fixed codes and that time-varying codes become better with increasing time period. In [10], an example is given where g is time varying. Here, we fix g and puncture in a time-varying manner.

One disadvantage of punctured codes is that error events at high rates and higher distances can be quite long. This was previously observed in [7]. To give an example, we have plotted in Fig. 3 the path length distribution of an $R = 4/5$ code with $M = 4$, $P = 4$. Paths at $d_{\text{free}} + 3 = 6$ can be up to 50 information bits long. For high rate codes, this requires a longer path memory.

IV. PERFORMANCE OF RCPC CODES ON GAUSSIAN AND FADING CHANNELS

The performance of RCPC codes on a Gaussian channel with soft decision and on interleaved Rayleigh and Rician channels using different degrees of channel state information has been evaluated using (9) and the distance spectrum $\{c_d\}$ in Tables I and II. For the Gaussian case simulation results are also given. Fig. 4 shows the transmission scheme over a nonfrequency-selective fading channel with multiplicative distortion a_F , with $E\{a_F^2\} = 1$ and

$$p_{a_F}(a_F) = 2a_F(1 + C/M) \exp\left(-\left(a_F^2(1 + C/M) + C/M\right)\right) \cdot I_0(2a_F\sqrt{C/M(1 + C/M)}). \quad (10)$$

For a Ricean channel, C/M is the ratio of the direct to diffusely reflected signal energy. With $C/M = 0$ we have Rayleigh fading, with $C/M \rightarrow \infty$ we obtain the Gaussian channel. In the fading case we assume perfect interleaving, which means that the CSI values a_F and the received code values y are statistically independent with density function (10). Since we consider a varying code rate, we describe the channel SNR by E_S/N_0 rather than by $E_b/N_0 = E_S/RN_0$. A reduced rate consequently means reduced throughput. The Viterbi algorithm for decoding the RCPC code uses the metric (3). In the following, we give the values for $\hat{a}_F = a_{UF}$ and $y = y_{ij}$ which the decoder uses in different cases, as well as the resulting P_d to be used in the evaluation of the BER in (9).

a) Gaussian channel [13]

$$\hat{a}_F = 1 \quad y \text{ analog or finely quantized}$$

$$P_d = \frac{1}{2} \operatorname{erfc} \sqrt{dE_S/N_0}.$$

TABLE I
PUNCTURING TABLES $a(l)$, c_d , AND a_d VALUES FOR RCPC CODES WITH MEMORY $M = 4$, PERIOD $P = 8$ AND RATES $R = P/(P + l) = 8/(8 + l)$, $l = 1, 2, 4, 6, \dots, 24$

RCPC CODE: $M = 4$ $P = 8$													
	8/9	8/10	8/12	8/14	8/16	8/18	8/20	8/22	8/24	8/26	8/28	8/30	8/32
	8/9	8/10	2/3	4/7	1/2	4/9	4/10	4/11	1/3	4/13	2/7	4/15	1/4
	1111 0111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111
	1000 1000	1000 1000	1010 1010	1110 1110	1111 1111	0000 0000	0000 0000	1000 1000	1100 1100	1110 1110	1111 1111	1111 1111	1111 1111
2	1										1000 1000	1010 1010	1110 1110
3	242	42											
4	4195	274	4										
5	63621	2658	0	2									
6	885316	21692	496	62									
7	11678199	154684	0	144	32	2							
8		1103834	10884	350	96	36	2						
9			0	2006	180	60	34	10					
10				5384	576	62	28	6					
11					1800	354	66	36	8	2			
12						4000	856	228	72	48	4		
13							354	114	72	56	20	2	
14								228	66	40	20	16	
15									104	36	36	36	32
16										256	104	24	0
17											56	18	0
18											184	74	32
19												48	
20												96	

	2	1											
	3	30	0										
	4	327	40	4									
	5	3430	274	0	2								
	6	37729	1686	108	18								
	7	408015	9842	0	32	16	2						
	8	534003	1280	74	24	14	2						
	9		0	308	32	18	14	4					
	10				608	80	22	10	6				
	11					296	76	20	14	8	2		
	12						544	164	54	22	16	4	
	13							78	30	24	20	12	2
	14								52	16	14	12	8
	15									24	12	12	16
	16									64	30	8	0
	17										16	14	0
	18										52	24	8
	19											16	
	20											32	

b) Rayleigh and Rice fading with hard decisions on y (YH) and no CSI, a not used (AN): YHAN

$$\hat{a}_F = 1, \quad y = \pm 1.$$

$$P_d = \begin{cases} \sum_{(d+1)/2}^d \binom{d}{e} P_0^e (1-P_0)^{d-e} & d \text{ odd} \\ P_{d-1} & d \text{ even} \end{cases} \quad (11)$$

with the channel bit error rate

$$P_0 = \int_0^\infty \frac{1}{2} \operatorname{erfc} \sqrt{a_F^2 \cdot E_s / N_0} P_{a_F}(a_F) da_F. \quad (12)$$

c) Rayleigh fading with soft decisions on y (YS) and full CSI (AS) with perfect estimation of the analogy amplitude a_F : YSAS

$$\hat{a}_F = a_F \quad y \text{ analog or finely quantized}$$

P_d is equivalent to the error probability of a d th order maximum ratio diversity system and can be upper bounded by [9]

$$P_d \leq \frac{1}{2} \left(\frac{1}{1 + E_s / N_0} \right)^d. \quad (13)$$

d) Rayleigh fading with hard decision (YH) and hard CSI

TABLE II
PUNCTURING TABLES $a(l)$ AND c_d VALUES FOR RCPC CODES WITH MEMORY $M = 3, 5, 6$.
PERIOD $P = 8$ AND RATES $R = P/(P+l) = 8/(8+l)$, $l = 1, 2, 4, 6, \dots, 14, 16$

RCPC CODE: $M = 3$									
	8/9	8/10	8/12	8/14	8/16	8/18	8/20	8/22	8/24
	8/9	4/5	4/6	4/7	1/2	4/9	2/5	4/11	1/3
	1110 1110	1110 1110	1110 1110	1110 1110	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111
	1001 0001	1001 1001	1001 1001	1101 1101	1101 1101	1111 1111	1111 1111	1111 1111	1111 1111
	0000 0000	0000 0000	0000 0000	0000 0000	0000 0000	1000 1000	1010 1010	1110 1110	1111 1111
2	21								
3	509	62	1						
4	7759	528	32	4					
5	69172	3894	156	0					
6	1168056	26580	736	160	18	4			
7	13088533	171784	2980	0	58	22	8		
8	1070980	12000	1808	144	34	16	10		
9		46688	0	392	118	24	20		
10				1040	286	92	22	48	
11				2664	580	200	58	0	
12					336	140		48	
13						284		0	
14							484		
15								0	
16									
17									
18									
19									
20									

(a)

RCPC CODE: $M = 5$									
	8/9	8/10	8/12	8/14	8/16	8/18	8/20	8/22	8/24
	8/9	4/5	2/3	4/7	1/2	4/9	2/5	4/11	1/3
	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111
	1000 0000	1000 1000	1010 1010	1110 1110	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111
	0000 0000	0000 0000	0000 0000	0000 0000	0000 0000	1000 1000	1010 1010	1110 1110	1111 1111
2	231								
3	231								
4	4229	80							
5	57612	782							
6	850398	8502	384	10					
7	11342400	54246	0	118					
8	151803233	426732	7616	316	16	4			
9		3239744	0	720	288	12			
10			143744	3048	256	100	8		
11			0	10438	496	322	100	2	
12					2656	234	144	32	
13					5608	798	232	134	8
14						368	148	64	
15						828	264	208	
16							384	160	
17								152	
18								496	
19									
20									

(b)

TABLE II (Continued)

RCPC CODE: M = 6									
	8/9	8/10	8/12	8/14	8/16	8/18	8/20	8/22	8/24
	1111 0111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111
	1000 1000	1000 1000	1010 1010	1110 1110	1111 1111	1111 1111	1111 1111	1111 1111	1111 1111
	0000 0000	0000 0000	0000 0000	0000 0000	0000 0000	1000 1000	1100 1100	1110 1110	1111 1111
2									
3	24								
4	740	24							
5	13321	376							
6	217761	3464	12						
7	3315491	30512	280	12					
8	48278177	242734	1140	74					
9		1800790	5104	388					
10			24640	1162	288	14			
11			106512	3542	0	82	8		
12				11666	1688	182	44	2	
13					0	320	80	16	
14					11232	810	138	70	8
15					0	1882	444	90	0
16						750	172	160	
17							474	0	
18								424	
19								0	
20									

(c)

TABLE III
COMPARISON OF RCPC CODES WITH PUNCTURED AND GENERAL CONVOLUTIONAL CODES

Rate	Memory	RCPC-Codes			Punctured Codes [8]			General K/N Codes	
		d _{free}	c _{d_{free}} /P	P	d _{free}	c _{d_{free}} /P	P	d _{free}	I
4/9	M = 3	4	5	2	4	5	2	4 [3]	-
		4	4	4					
2/3	M = 6	6	1.5	2	6	1.5	2	7 [3]	-
		4	4	4					
4/5	M = 3	3	2.2	4	3	3.5	4	2 [12]	-
		4	3	4	4	3	4	-	
8/9	M = 3	2	2.6	8	2	0.8	8	-	-
		3	3	8	3	1.6	8	-	
4/7	M = 3	4	0.5	4	-	-	-	4 [12]	-
		6	1.25	4	-	-	-	6 [12]	

* With memory to give an equal number of total branches per information bit in the VA.

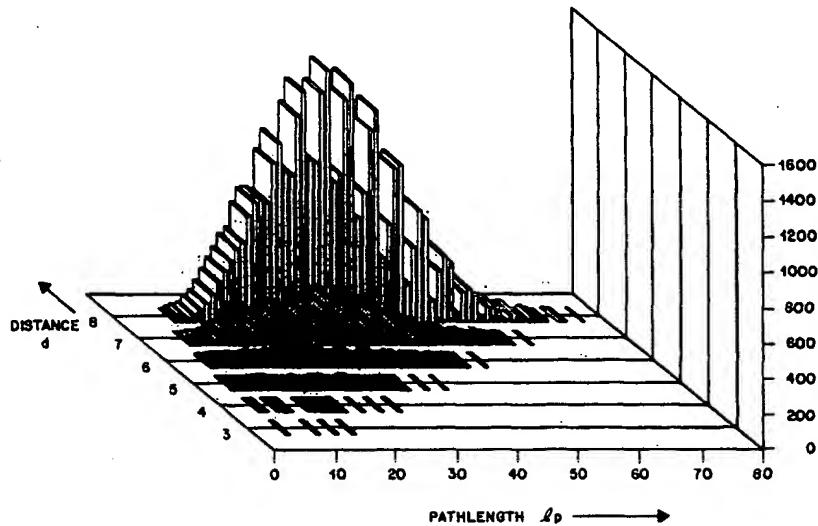


Fig. 3. Number of paths with length l_p at distance d for a RCPC code of rate $R = 4/5$, $P = 4$, $M = 4$.

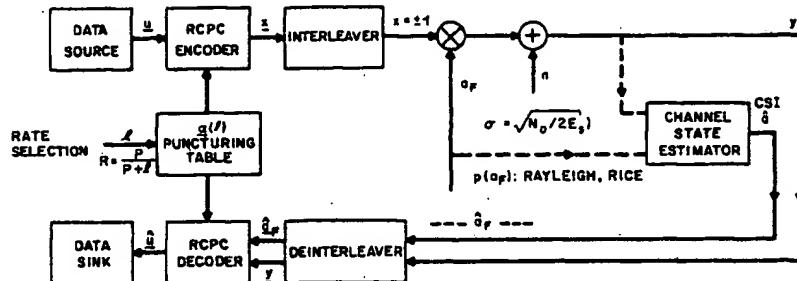


Fig. 4. Coded data transmission using RCPC codes with interleaving and channel state information (CSI) on fading channels.

(AH): YHAM

$$\hat{a}_P = \begin{cases} \log \frac{1 - P_B}{P_B} & \text{for } a_P < a_T \\ \log \frac{1 - P_G}{P_G} & \text{for } a_P \geq a_T \end{cases} \quad (14)$$

where a_T is an optimized threshold. By partial integration in (12), we obtain P_G as the channel BER above the threshold a_T and P_B as the BER below a_T [9]. P_B can be calculated in closed form. The result is given in [9] and [14].

Fig. 5 shows the simulated RCPC performance for the Gaussian channel. Simulation and an analytical upper bound at a BER of 10^{-5} differ only by approximately 0.2 dB. The performance curves for the Rayleigh channel with the scheme described in Fig. 4 and in Section IV show that RCPC codes with rates between $4/5$ and $1/3$ cover a broad range of SNR and BER. At a BER of 10^{-5} a 1-bit CSI gains 7.9 dB at rate $2/3$ and 2.6 dB at rate $1/3$ relative to hard decisions without CSI. The Rice channel with $C/M = 7$ dB is typical for mobile satellite communications to ships with small antennas, to cars in open areas, and to airplanes with hemispherical antennas [14]. As Fig. 7 shows, the RCPC codes using basically the same encoder and the same Viterbi decoder with only 16 states and two branches per state can cover a SNR between 4

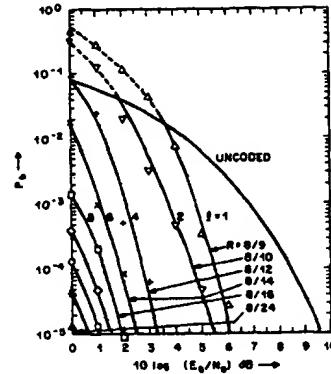


Fig. 5. BER performance of RCPC codes on Gaussian channels, $N = 3$, $P = 8$, $l = 1, 2, 4, \dots, 16$, rate $= 8/(8+l)$, $M = 4$: simulation with soft decision.

and 17 dB to achieve a BER of 10^{-5} . This means that the same link can serve users with a gain variation of 13 dB by adaptively changing the rate between $4/5$ and $1/3$ at the expense of throughput.

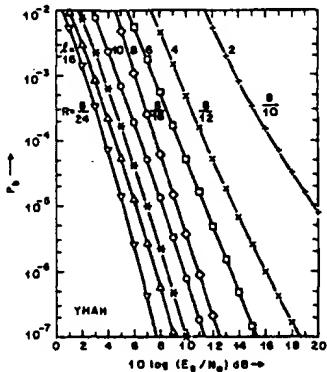


Fig. 6. BER performance of RCPC codes on an interleaved Rayleigh channel, $N = 3$, $P = 8$, $l = 2, 4, \dots, 16$, rate $R = 8/(8+l)$, $M = 4$; analytic upper bound YHAN; hard decision, 1 bit CSI.

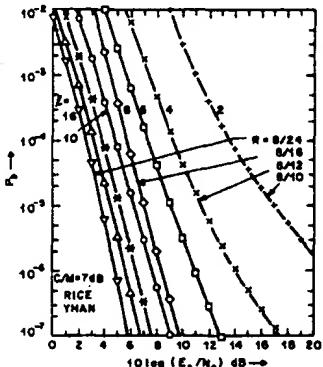


Fig. 7. BER performance of RCPC codes on an interleaved Rice channel $C/M = 7$ dB, YHAN: hard decision, no CSI, $N = 3$, $P = 8$, $l = 2, 4, \dots, 16$, rate $R = 8/(8+l)$, $M = 4$, analytic upper bound.

V. THE APPLICATION OF RCPC CODES TO HYBRID ARQ/FEC TRANSMISSION

For data transmission schemes where a return channel is available, automatic repeat request (ARQ) schemes guarantee high reliability at high and moderate channel qualities. The throughput performance degrades if the channel is bad or time varying. A combination with forward error correction (FEC) to form a hybrid system combines the advantages of both methods. A broad variety of hybrid systems has been suggested, a good overview of the literature is given in [3] and [4]. Some more recent proposals can be found in [15]-[19]. A so-called "type II hybrid system" [4] uses a rate 1/2 invertible code and alternately sends coded sequences ("information" and "parity" sequences) which are either detected to be correct or combined for FEC decoding. Block and convolutional FEC codes have been used and extensions to rate 2/3 and 3/4 have been reported [4]. Typically, the "throughput-versus-channel quality"-curves exhibit a saddle at the FEC code rate and some of the schemes require different decoders at different stages of the detection and decoding algorithm. A more natural scheme than alternate code and parity transmission seems to be successive parity transmission to build up a code which is finally powerful enough to decode the message. Consequently, none of the already transmitted code bits are thrown away but are used to improve FEC decoding, preferably with ML decoding. As in most other hybrid schemes an outer high rate block-code detects decoding errors.

This method goes back to Mandelbaum [5] and includes variations like code combining [15], [16] and memory ARQ [17].

We propose to use rate compatible punctured convolutional codes (RCPC Codes) with Viterbi algorithm (VA) decoding for successive parity transmission. The principle of the proposed ARQ/FEC scheme is not to repeat information or parity bits if the transmission is unsuccessful as in previous type II ARQ/FEC schemes, but to transmit additional code bits of a lower rate RCPC code until the code is powerful enough to enable decoding. This includes several decoding attempts on the receive side, which seem especially feasible for applications in mobile or low rate systems where the data rate is in the kbit/s range and the VA decoder would run on a chip at a higher rate.

A. ARQ/FEC Protocol with RCPC Codes

The ARQ/FEC protocol performs the following steps while transmitting a block I of information, which contains typically $n = 100$ to 1000 bits.

Encoding:

- 1) Add n_c parity bits to form an error detection code C_0 , i.e., 16 or more CRC bits.
- 2) Add M "0" or known bits to properly terminate the encoder memory and the decoder trellis for the "information" block $(n + n_c)$. M is the memory of the convolutional encoder.
- 3) Use the $1/N$ encoder to encode the convolutional code and store it at the transmitter, possibly in a matrix as in Fig. 8. The matrix is only conceptual and can be modified in several ways to accommodate interleaving.

Transmitting and decoding:

- 1) Transmit the first P columns of the matrix.
- 2) Transmit additional columns of the matrix up to index l according to the puncturing table.
- 3) Decode RCPC code of rate $P/(P + l)$ with Viterbi algorithm.

- 4) Check the syndrome of code C_0 . If the syndrome is zero output I and send ACK to the transmitter. If the syndrome of C_0 is not zero increase l and repeat steps (2)-(4).

If decoding is still not successful for $l = (N - 1)P$, i.e., $R = 1/N$, several possibilities exists. A higher order protocol could take over or ask for a repetition of the whole procedure starting at step 1). Code combining as suggested by Chase [15] could be used to repeat the whole $1/N$ code L , times and combine it with the already transmitted code. Actually, the VA uses in this case the same trellis as for all the RCPC codes, but the ML metric (3) is changed to

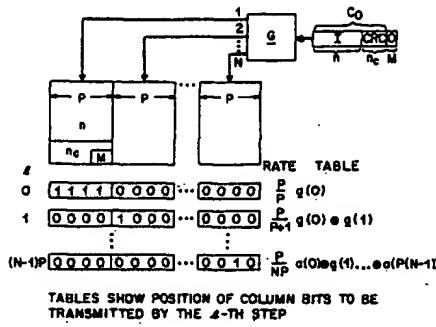
$$\lambda_j = \sum_{r=1}^{L_r} \sum_{i=1}^N a_{ij} a_{ijr} x_{ij} y_{ijr} = \sum_{i=1}^N a_{ij} x_{ij} \sum_{r=1}^{L_r} a_{ijr} y_{ijr} \quad (15)$$

where index r indicates the r th repeated transmission. Note that by repeating all code bits L , times the free distance is increased by a factor of L , [15] and the numbers of the weight spectrum are unchanged

$$c_{d_r} = c_d \quad d_r = L_r \cdot d \text{ for } d = d_{\text{free}} + i.$$

Note that a repetition code is almost optimal only for low rates such as 1/8, 1/16 [15]. By halving the code rate the distance usually is more than twice as shown in Table I for memory $M = 4$. This enables a gain in E_b/N_0 on the Gaussian channel. This type of code combining as expressed in (15) can be viewed as maximum ratio combining of the received y_{ijr} values. The combined y values which equal the inner sum of the right side in (15) are then used by the original code for ML decoding.

In the following analysis, we will assume that decoding stops with the $1/N$ code in step 4). Consequently, there exists



a probability that the frame with information block I was not error-free decoded. This frame error rate FER will be calculated and should be sufficiently small. It should be pointed out that we are completely free in which steps we increase I . The number of steps depends on the ratio of frame arriving time (information rate) to frame decoding time (decoder speed). For extremely varying channel conditions an exponential increase could be useful, i.e., $I = 1, 2, 4, 8, 16$. This would mean to perform less FEC decoding attempts than the maximum number possible. This is at the cost of a slightly smaller throughput because in the example, the 10th attempt could have been successful, saving the transmission 11–16 in the above example.

With this type of protocol an acknowledgement (ACK) is only sent after successful decoding of a frame. The encoder would send more and more code bits until it receives the ACK. Assume that decoding is successful at step I_S . If there is a round-trip delay T_D involved in the transmission with rate $1/T_S$, the encoder would transmit T_D/T_S unnecessary bits, which would slightly reduce the throughput. This will be shown in the analysis. If the ACK is lost, the encoder keeps transmitting unnecessary code bits until $I = (N - 1)P$. If the probability of an ACK loss or error is small the effect on the throughput is negligible. Thus, the return channel is used only once per frame and a bad return channel does not confuse the basic protocol.

B. Throughput Analysis

We will describe the performance of the hybrid ARQ/FEC scheme by two figures of merit: the average throughput R_{AV} and the probability that a frame or packet with n information bits is finally not correctly decoded, the so-called frame error rate FER. The throughput is

$$R_{AV} = \frac{n}{n + n_c + M} \cdot \frac{P}{P + I_{AV}}. \quad (16)$$

I_{AV} is the average number of additionally transmitted bits per P information bits. From Fig. 8 this can be interpreted as the average number of additionally transmitted columns. $P/(P + I_{AV})$ is the effective code rate of the FEC code. Due to the overhead of n_c parity check bits and M terminating bits with $n_c + M \leq n$, the throughput is slightly smaller than the effective FEC code rate. We assume that the error detecting code C_0 achieves a very low probability of undetected errors which will be ignored in the subsequent analysis. If the $(n + n_c, n)$ error detection code C_0 falls into the class which satisfies [20]

$$\text{Prob}\{\text{error undetected}\} < 2^{-n_c}$$

a very low error probability can be achieved when n_c is at least 16. Typically, one would use a cyclic redundancy check CRC common in data transmission protocols. Let I_k be the

sequence number of the decoding attempt using the RCPC code with puncturing tables $a(I_k)$ with

$$I_1 < I_2 < \dots < I_k < \dots < I_K.$$

Let $P_{EF}(I_k)$ be the probability that the FEC decoding at the k th step results in errors which are detected by C_0 . We will give an approximate analysis where we assume that the K decoding attempts have statistically independent outcomes. Then we have

$$I_{AV} = \sum_{k=1}^K I_k (1 - P_{EF}(I_k)) \cdot \prod_{i=0}^{k-1} P_{EF}(I_i) + I_K \prod_{i=0}^K P_{EF}(I_i) \quad (17)$$

where we define $P_{EF}(I_0) = 1$.

The frame error rate of the unsuccessful ARQ/FEC procedure is then

$$\text{FER} = P_{EF}(I_K). \quad (18)$$

We can include in the analysis a delay T_D caused by the round-trip time and the decoding time by replacing I_k in (17) by

$$I_k := \min\{I_K, I_k + PT_D / (T_S(n + n_c + M))\}. \quad (19)$$

Similarly, a loss or error in the ACK signal on the return channel, causing unnecessary transmission up to I_K , can be accommodated. For the probability $P_{EF}(I_k)$, we only can give an upper bound by using (9) to calculate the probability that an error event occurs for the transmitted bits

$$P_{EF}(I_k) < 1 - \left(1 - \frac{1}{P} \sum_{d=d_{\text{thr}}}^{\infty} a_d(I_k) \cdot P_d\right)^{n+n_c+M}. \quad (20)$$

The values a_d depend on the code, P_d depends on the channel and the detection scheme and can be selected from the values given in Section IV and in [9]. Recall that P_d usually assumes sufficient interleaving. By using (20) in (17) and (16), we obtain a lower bound for the throughput R_{AV} and by using (20) in (19), we get an upper bound on the FER.

Two examples will be given to show the performance of the RCPC codes in the hybrid ARQ/FEC scheme. The same code with $M = 4, N = 3, P = 8$ will be used, which has fairly low decoding complexity with 16 states and 32 add, compare and select operations per decoded bit. Fig. 9 shows the throughput on a Gaussian channel with hard and soft decisions as calculated by the lower bound method described. The simulations show that for this case the bound is accurate within 1 dB. The benefits of soft decisions are clearly observable. Due to the overhead R_{AV} approaches 0.97 for high SNR. At low SNR, R_{AV} approaches 0.323 and the frame error rate is high. The RCPC scheme which uses adaptive code rates between 1 and 1/3 has generally a better throughput than comparable schemes described in [4] and its throughput curve constitutes a kind of envelope to the corresponding fixed rate schemes. In Fig. 10 the performance on a fully interleaved Rayleigh channel is given. In this case, decoding starts with $I = 1$, i.e., the 8/9 code and ends with the 1/3 code. Several degrees of CSI are given as described in Section IV: YHAN, YHAH, and YSAS. Note that for a Rayleigh channel with $E_s/N_0 = 10$ dB a frame error rate of less than 10^{-6} and a throughput of 0.7 can be easily achieved. Simulation results and analytic bounds agree within 1 dB.

VI. THE APPLICATION OF RCPC CODES TO UNEQUAL ERROR PROTECTION

In many applications it is required to provide different levels of error protection for different parts of an information sequence or block. Some examples have been given in the introduction. We assume that the source or the source coder supplies us with some information about the relative impor-

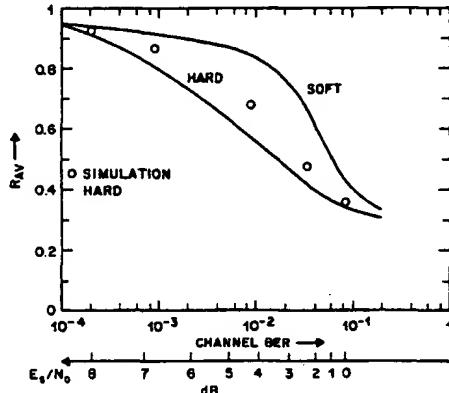


Fig. 9. ARQ/FEC throughput with RCPC codes on a Gaussian channel with soft and hard decisions (BSC). Code: $P = 8, N = 3, l = 0, 1, 2, 4, \dots, 16, M = 4$. Frame: $n = 8 \times 125, n_c = 28$.

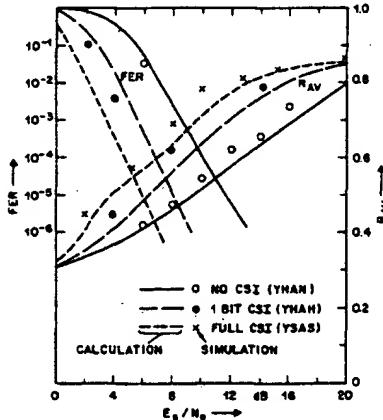


Fig. 10. ARQ/FEC throughput and frame error rate FER on an interleaved Rayleigh channel, hard decision, and CSI. Code: $P = 8, N = 3, l = 1, 2, 4, \dots, 16, M = 4$. Frame: $n = 8 \times 48, n_c = 34$.

tance or susceptibility to errors of certain information bits or groups of information bits. We will call this source significance information (SSI). An example of an SSI would be the required BER after decoding for certain groups of information bits. Assume, that in a block of n information bits we have K groups with n_k information bits in the k th group requiring an BER of P_{bk} after decoding. In this case, P_{bk} would be the SSI and

$$\sum_{k=1}^K n_k = n.$$

An example is given in Fig. 11 where we have ordered the information bits according to their relative importance and their error protection needs P_{bk} . Of course one could now separately encode the K groups by K different encoders and use K different decoders according to the protection needs. Besides the increased overhead and complexity some numbers n_k might be rather small and exclude the use of powerful longer codes. Instead, we wish to use *one* encoder and *one* decoder which provides the protection requirements with a minimum of redundancy and overhead. Very little is known in the literature about unequal error protection of this type by block or convolutional codes. A recent reference is [21] where more literature is quoted. Most recently the extension of this

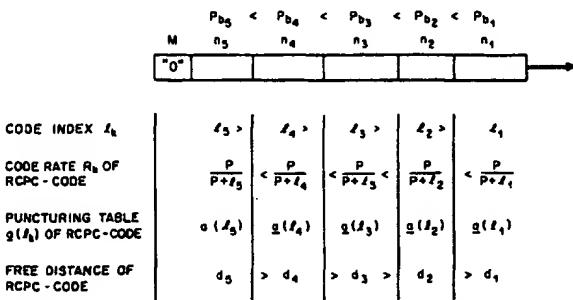


Fig. 11. Data frame with information bits grouped according to their error protection requirement (source significance information P_{bk}). RCPC codes assigned to the groups.

concept to convolutional codes has been reported [26]. It seems, however, that only a limited amount of variability can be introduced and only short or low rate codes have been used. Furthermore, no ML decoding results have been shown. Another interesting unequal protection scheme with two codes has been reported in [22], which combines block and convolutional codes.

RCPC codes are well suited for this application. In the example of Fig. 11, the ordered information bits are shifted into the shift register of a $1/N$, memory M RCPC code. During the n_1 information bits the puncturing matrix $\alpha(l_1)$ is used as the relevant rule for the multiplexer. As soon as the first bit of the second group enters the encoder the puncturing table, $\alpha(l_2)$ will be used. After another n_2 information bits or encoder shifts, the table is switched to $\alpha(l_3)$, etc. The procedure is easier to follow when n_k is an integer multiple of the puncturing period P , however, this is not a strict requirement. At the k th step $P + l_k$ code bits are transmitted per P information bits using the RCPC code with index l_k and rate $R_k = P/(P + l_k)$. The frame is terminated after the group n_K by shifting M '0' bits into the shift register, thus transmitting M/R_K overhead code bits necessary for proper termination of the trellis. The average code rate is then

$$R = \frac{\sum_{k=1}^K n_k}{\sum_{k=1}^K n_k \cdot (P + l_k)/P + M(P + l_k)/P}$$

The rate-compatibility condition (7a) is crucial. In a transitional phase between two matrices $\alpha(l_k)$ and $\alpha(l_{k+1})$ where $l_{k+1} > l_k$ we have to guarantee, that despite the transition, the distance properties of all paths originating in code l_k do not suffer a loss of distance due to transitions, thus guaranteeing at least the designed performance. Fig. 12 shows the same error paths relative to the correct all-zero path in RCPC-code l_k : path 1 having distance d_1 is not effected by a transition, whereas path 2 extends into the lower rate code l_{k+1} . The rate-compatibility condition (7a) guarantees that the l_{k+1} code does not puncture any '1's on the path. Due to less puncturing it might happen that additional '1's occur on the part B of the path, which would cause a $d_2 > d_1$. Therefore, some of the $\{a_{ij}\}$ and $\{c_{ij}\}$ values have to be assigned to higher distances if the path is not terminated at the transition. Thus, the rate compatibility restriction guarantees that transitional paths do have a distance which is at least the distance of the same path within the higher rate code and at most the distance of the same path within the lower rate code. If two punctured codes without the rate compatibility restriction are used at a rate transition, it can happen that a transitional path has a distance which is even lower than the distance of the same path

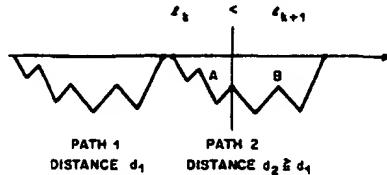


Fig. 12. Error paths in a time varying RCPC code during a transition period between code rates.

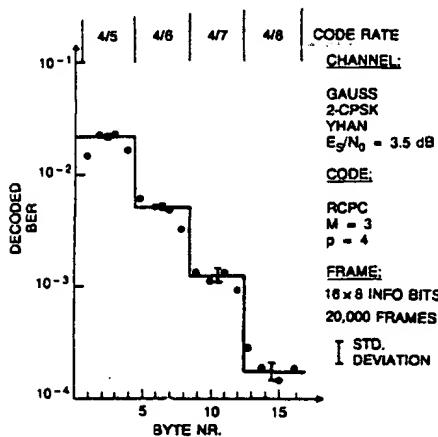


Fig. 13. Measured short term bit error rate in a data frame with unequal error protection using RCPC codes.

within the higher rate code. This would lead to a bad error behavior in the transition region. One can easily construct such an example.

It is useful to order the information as described with increasing importance. Decoding errors typically occur at the beginning of paths and the last M information bits of each error path are error free. However, the transitional paths (path 2) to lower rates generally have higher distances and therefore errors occur with lower probability. Simulations show that this type of ordering leads to a smaller number of total errors. Error paths can extend over many transitions. All paths will be properly terminated by the tail forcing the decoder to the all-zero state. Ordering the information bits from lower to higher rate codes would require proper zero termination at each change of rate. We conclude this section by giving an example of the simulated performance of the RCPC code with unequal error protection within one data frame of 128 bits. As Fig. 13 shows the code rate is changing from 4/5 to 4/8 while the measured short term BER is improving by a factor of 100. As discussed above, the bits close to a transition to the next lower rate get a better protection.

VII. CONCLUSIONS

We have described the concept of forward error correction codes which are obtained from low rate convolutional codes by puncturing them periodically in a rate-compatible manner. Surprisingly, this restriction still produced good codes relative to the best known codes of comparable complexity. In such a way, a whole family of codes with different rates is available using the same encoder and the same ML-decoder employing the Viterbi algorithm. Only the puncturing rule, i.e., the multiplexer switching pattern is changed. Since in most transmission schemes, channel rate is fixed due to modulation and channel filter requirements, changing the code rate means a change of information rate and a buffer for the incoming

information stream is required. One disadvantage of punctured codes with high rate is that error paths can be quite long, as shown in Fig. 3 and in [7]. This requires either long path memories in the VA, framed transmission as employed in most packet oriented transmission schemes, or periodically inserted synchronization patterns or known data bits. In a decoder implementation, the basic Viterbi algorithm runs on the same trellis for all the codes. This suggests a VLSI for the Viterbi-decoding algorithm where only the control and the memory access is changed according to the rate, the channel conditions, and the channel state information (CSI).

Since the compatibility condition ensures that only incremental redundancy is used for lower rate codes, RCPC codes are suitable for ARQ/FEC protocols and unequal error protection as shown in Sections V and VI. Of course, both methods can be combined to adapt unequal error protection to varying channel conditions. For instance, the transmission of a frame with a set of unequal error protection codes using rates (1, 8/10, 8/14) can be upgraded to a set of (8/10, 8/12, 8/16) codes when the channel conditions get worse by transmitting another two check bits per 8 information bits.

The concept of RCPC codes can be combined with the concept of nested codes [25] where the Viterbi decoder can work also with $M = 1, M = 2, \dots$ on a reduced set of states. Codes have been found which satisfy both the RCPC and the nested condition. Furthermore, the concept of RCPC codes could also be used to find a family of "good" codes with high memory suitable for sequential decoding [3]. "Good" would mean that all the codes with different rates need a rapidly growing distance profile [3]. In an ARQ/FEC application, the error detection via the C_0 code could be replaced by a time-out condition in the Fano or stack algorithm.

The design of combined source and channel coding as described for voice transmission in [23] and for image transmission in [24] gets a new degree of freedom by using RCPC codes because the code rate can be changed within a data frame and during transmission according to the needs [27]. In other words, source significance information SSI, channel state information CSI, and check information from the decoded bits can be used for the adaptation of the encoding and ML-decoding algorithm of RCPC codes.

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